## EE 261 The Fourier Transform and its Applications Fall 2003, Handout 41 Midterm Solutions

1. The signal f(t) is shown below. Its Fourier transform is F(s).



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That's given by f(-t), or, without writing the variable,  $f^-$ . Its Fourier transform is  $F^-$ , or, writing the variable, F(-s).

The next one is



To express this in terms of f we first reverse f to  $f^-$  (which gives the preceding picture) and then shift to the right by 2, giving  $f^-(t-2)$ . The Fourier transform of this is  $e^{-2\pi i 2s} \mathcal{F}(f^-)(s) = e^{-4\pi i s} (\mathcal{F}f)^-(s) = e^{-4\pi i s} F(-s)$ .

And next we have



That's just f(t-1), and the Fourier transform is  $e^{-2\pi i s}F(s)$ .



This is a little trickier. The figure is twice as large in the horizontal direction, which is accomplished by forming f(t/2) and also scaled up by 2 in the vertical direction, so that makes 2f(x/2). The combination of the two scalings gives a curve of the same shape, just twice as large. Leaving either of the two scalings out would distort the shape.

The Fourier transform is

$$2 \times \frac{1}{\frac{1}{2}}F(2s) = 4F(2s).$$

The traffic picks up with:



This is given by f(t) + f(t+2) so its Fourier transform is  $F(s) + e^{4\pi i s}F(s) = (1 + e^{4\pi i s})F(s)$ .

Finally, heading off in opposite directions:



That's just f(t) + f(-t) so the Fourier transform is F(s) + F(-s)

2. (15 points) Recall that the inner product of two functions in  $L^2([0,1])$  is defined by

$$(f,g) = \int_0^1 f(t)\overline{g(t)} dt.$$

Let  $\{\varphi_n(t)\}\$  be an orthonormal basis for  $L^2([0,1])$ . Remember that orthonormality means

$$(\varphi_n, \varphi_m) = \begin{cases} 1, & , & n = m \\ 0, & n \neq m \end{cases}$$

To say that the  $\{\varphi_n\}$  form a basis means we can write any function f(t) in  $L^2([0,1])$  as

$$f(t) = \sum_{n=1}^{\infty} (f, \varphi_n) \varphi_n(t).$$

Define a function of two variables

$$K(t,\tau) = \sum_{n=1}^{\infty} \varphi_n(t) \overline{\varphi_n(\tau)}$$

Show that

$$(f(t), K(t, \tau)) = f(\tau).$$

(Ignore all questions of convergence of series, *etc.* Because of this property,  $K(t, \tau)$  is called a *reproducing kernel*; taking the inner product of f(t) with  $K(t, \tau)$  'reproduces' the value of f at  $\tau$ . In Electrical Engineering literature sampling theorems are often expressed in terms of reproducing kernels.)

We compute the inner product of f(t) and  $K(t, \tau)$ . You can write this out in terms of integrals or you can use simply the properties of (complex) inner products. I'll show you both approaches.

$$(f(t), K(t, \tau)) = \left( f(t), \sum_{n=1}^{\infty} \varphi_n(t) \overline{\varphi_n(\tau)} \right)$$
$$= \sum_{n=1}^{\infty} (f(t), \varphi_n(t) \overline{\varphi_n(\tau)})$$
$$= \sum_{n=1}^{\infty} \varphi_n(\tau) (f(t), \varphi_n(t))$$

(The inner product is 'with respect to t', and so the  $\overline{\varphi_n(\tau)}$  are constants in each inner product. Being in the second slot, they come out with a conjugate, *i.e.* as  $\varphi_n(\tau)$ )

But this last expression is exactly how to write  $f(\tau)$  in terms of the orthonormal basis,

$$\sum_{n=1}^{\infty} (f, \varphi_n) \varphi_n(\tau) = f(\tau).$$

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If you had trouble keeping your t's and  $\tau$ 's straight in what we just did, here's how the calculation looks with integrals.

$$(f(t), K(t, \tau)) = \int_0^1 f(t) \overline{K(t, \tau)} dt$$
  
$$= \int_0^1 f(t) \overline{\left(\sum_{k=1}^\infty \varphi_n(t) \overline{\varphi_n(\tau)}\right)} dt$$
  
$$= \int_0^1 f(t) \left(\sum_{k=1}^\infty \overline{\varphi_n(t)} \varphi_n(\tau)\right) dt$$
  
$$= \sum_{k=1}^\infty \left(\int_0^1 f(t) \overline{\varphi_n(t)} dt\right) \varphi_n(\tau)$$
  
$$= \sum_{k=1}^\infty (f, \varphi_n) \varphi(\tau)$$
  
$$= f(\tau)$$

3(a) (10 points) Find the Fourier transform of the signal shown below.



The signal is made out of the unit step function H(t), so for sure we'll need to use

$$\mathcal{F}H(s) = \frac{1}{2}(\delta + \frac{1}{\pi i s}).$$

The part of the signal on the right is just a shift of H(t) to the right, and that's H(t-1). To get the part on the left we first reverse to  $H^-$ , then shift to the left, making  $H^-(t+1)$  (which you could also write as H(-(t+1)) = H(-t-1), but I won't) and then flip it upside down, to  $-H^-(t+1)$ . The whole signal is then

$$f(t) = H(t-1) - H^{-}(t+1)$$

The Fourier transform of this is

$$\begin{aligned} \mathcal{F}f(s) &= e^{-2\pi i s} \mathcal{F}H(s) - e^{2\pi i s} (\mathcal{F}H^{-})(s) \\ &= e^{-2\pi i s} \mathcal{F}H(s) - e^{2\pi i s} (\mathcal{F}H)^{-}(s) \\ &= e^{-2\pi i s} \mathcal{F}H(s) - e^{2\pi i s} \mathcal{F}H(-s) \\ &= e^{-2\pi i s} (\frac{1}{2}\delta(s) + \frac{1}{2\pi i s}) - e^{2\pi i s} (\frac{1}{2}\delta(-s) + \frac{1}{2\pi i (-s)}) \\ &= \frac{1}{2} (e^{-2\pi i s} - e^{2\pi i s})\delta(s) + \frac{1}{2\pi i s} (e^{-2\pi i s} + e^{2\pi i s}) \\ &= -i(\sin 2\pi s)\delta(s) + \frac{1}{\pi i s} \cos 2\pi s \\ &= \frac{1}{\pi i s} \cos 2\pi s \end{aligned}$$

As a check, notice that the Fourier transform is purely imaginary and odd, which it should be.

3(b) (10 points) Recall the triangle function

$$\Lambda(t) = \begin{cases} 1 - |t|, & |t| < 1\\ 0, & \text{otherwise} \end{cases}$$

and consider the signal

$$f(t) = 1 + \Lambda(3t) * III_{1/3}(t).$$

Sketch the graph of f(t), find the Fourier transform  $\mathcal{F}f(t)$ , and sketch its graph. Comment on what you see.

Taking the convolution  $\Lambda(3t) * III_{1/3}(t)$  produces the sum

$$\Lambda(3t) * \operatorname{III}_{1/3}(t) = \sum_{k=-\infty}^{\infty} \Lambda(3(t-\frac{k}{3})) = \sum_{k=-\infty}^{\infty} \Lambda(3t-k)$$

A plot of the shifted triangles from k = -4 to 4 looks like this:



If you draw your graph carefully enough (or actually do the algebra) you'll see that the overlaps are such that when you *add* the shifted triangles you get a plot that looks like:



And you would have concluded that  $\Lambda(3t) * \prod_{1/3}(t) = 1$ , and that f(t) = 2, and, so, for that matter that  $\mathcal{F}f(s) = 2\delta$ 

However, if you didn't draw the graph carefully enough you had a second chance to come to the same conclusion by taking the Fourier transform:

$$\mathcal{F}f = \delta + \left(\frac{1}{3}\mathrm{sinc}^2(\frac{1}{3}s)\right)(3\mathrm{III}_3(s))$$
$$= \delta + \sum_{k=-\infty}^{\infty}\mathrm{sinc}^2(\frac{k}{3})\delta(s-3k)$$

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Now the thing you have to realize is that  $\operatorname{sinc}^2(k/3)$  is 1 at k = 0 and is zero at the points where k/3 is an integer, *i.e.* where k is a multiple of 3. The zeros of  $\operatorname{sinc}^2$  line up with the impulses  $\delta(s-3k), k \neq 0$  and hence kill them off. What remains is

$$\mathcal{F}f = \delta + \delta = 2\delta.$$

I won't plot this. But having seen it, if you didn't get the graph of f(t) right you should now see that it's just  $2\mathcal{F}^{-1}\delta = 2$ .

3(c) (10 points) As one of our applications of Fourier transforms and convolutions to differential equations (Lecture 8) we showed that the ordinary differential equation

$$u'' - u = f$$

has solution

$$u(t) = \frac{1}{2} \int_{-\infty}^{\infty} e^{-|t-\tau|} f(\tau) \, d\tau.$$

We know from elementary courses that the general solution of the equation should include a solution of the homogeneous equation u'' - u = 0, and so be of the form

$$u(t) = c_1 e^t + c_2 e^{-t} + \frac{1}{2} \int_{-\infty}^{\infty} e^{-|t-\tau|} f(\tau) \, d\tau.$$

A student sent an email pointing this out and wondering why methods based on the Fourier transform will not produce such a solution. Why do you think that is? (You *need not* look back at Lecture 8.)

The reason is that neither  $e^t$  nor  $e^{-t}$  have Fourier transforms;  $e^t$  grows too fast at  $+\infty$  and  $e^{-t}$  grows too fast at  $-\infty$ . No argument based on using the Fourier transform can handle these functions as solutions to the differential equation.

4. (25 points) Let f(t) be a band-limited signal whose Fourier transform  $\mathcal{F}f(s)$  is shown below. The spectrum is nonzero only for 2 < |s| < 3.



According to the sampling theorem, we can reconstruct this signal by a sinc interpolation with sample points taken at a rate of 6 Hz. Since the spectrum is concentrated in two islands, effectively with what one might say is a bandwidth of 1 Hz each, it would seem we should be able to do better. Arguing as in the derivation of the sampling theorem, find an interpolation formula that uses a sampling rate of 2 Hz. Can you do better still?

If we convolve  $\mathcal{F}f$  with III<sub>2</sub> each 'spectral island' is shifted enough so that there is no overlap and the copies of the islands are still distinct when added up. The (partial) picture (squeezed slightly) is:



We still have to recover the original Fourier transform (and then take the inverse). To do that we multiply by the sum of *two*  $\Pi$ 's of width 1, one centered at 2.5 and one centered at -2.5. The picture is:



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If we now take the inverse Fourier transform we get

$$f(t) = (2\cos 5\pi t \operatorname{sinc} t) * (\frac{1}{2} \operatorname{III}_{1/2}(t) f(t))$$
  
=  $(\cos 5\pi t \operatorname{sinc} t) * \left(\sum_{k=-\infty}^{\infty} f(k/2)\delta(t-k/2)\right)$   
=  $\sum_{k=-\infty}^{\infty} f(k/2)(\cos 5\pi t \operatorname{sinc} t) * \delta(t-k/2)$   
=  $\sum_{k=-\infty}^{\infty} f(k/2)\cos(5\pi(t-k/2))\operatorname{sinc}(t-k/2)$ 

We also see that if we tried to periodize  $\mathcal{F}f$  with  $\Pi_p$  for any p < 2 then the islands would overlap. We could not recover  $\mathcal{F}f$  by cutting off, nor could we then recover f.